

Amendments to the Specification:

In the English translation document, please delete the term --Description-- at page 1 line 1 before the title.

In the English translation document, please add the section heading and paragraph at page 1 line 6, after the title, as follows:

--CROSS REFERENCE TO RELATED APPLICATIONS

This application is the US National Stage of International Application No. PCT/EP2005/050532, filed February 8, 2005 and claims the benefit thereof. The International Application claims the benefits of German application No. 102004006756.2 DE filed February 11, 2004, both of the applications are incorporated by reference herein in their entirety.--

In the English translation document, please add the section heading and paragraph at page 1 line 6, after the newly added CROSS REFERENCE TO RELATED APPLICATIONS section, as follows:

--FIELD OF INVENTION

The present invention relates to setting up a packet-oriented multimedia connection using an Interactive Voice Response System.--

In the English translation document, please add the section heading at page 1 line 6, after the newly added FIELD OF INVENTION section, as follows:

--BACKGROUND OF THE INVENTION--

In the English translation document, please add the section heading at page 9 line 21, as follows:

--SUMMARY OF INVENTION--

In the English translation document, please add the paragraphs at page 9 line 29, as follows:

The document US 6,512,818 discloses a method, in which a continuous connection is set up between a first and second subscriber using a Voice Response Unit (VRU), with a first

connection being set up first between the first subscriber and the VRU and a second connection being set up between the VRU and the second subscriber, which are then switched together by the VRU to form the continuous connection such that the VRU is no longer part of the switched connection. Ringback to the first subscriber has no response.

The document US 6,574,335 discloses a method, in which a first subscriber, linked via a line-oriented connection to a packet-oriented network, sets up a first connection to a second subscriber. Ringing at the second subscriber is displayed by the second subscriber to the first subscriber with the aid of a ringback message, which is necessary because a line-oriented ringback due to the existing line-oriented connection is not possible. In view of this message, the first subscriber selects a ringback tone that is suitable for the network of the second subscriber from a set of previously stored ringback tones.

If the first connection is redirected to a further subscriber via a gate controller, a second connection is set up to the further subscriber and a further ringback message is sent from there to the initiator of the second connection as part of the setting up of the second connection. Notification of this further ringback to the first subscriber, with which the already accepted first connection still exists, has no response.

In the English translation document, please amend the paragraph at page 12 lines 1-2, as follows:

Further advantageous embodiments of the invention will emerge from the ~~subordinate or independent~~ claims.

In the English translation document, please add the section heading at page 12 line 30, as follows:

--BRIEF DESCRIPTION OF THE DRAWING--

In the English translation document, please add the section heading at page 13 line 5, as follows:

--DETAILED DESCRIPTION OF INVENTION--

In the English translation document, please amend the paragraph at page 14 line 33-page 15 line 13, as follows:

First of all a first connection TDM, RTP/RTCP_{A/IVR} is set up between the subscriber A and the system IVR, which is assigned to the packet-oriented network IN_{IVR}. Because the subscriber A is assigned to the line-oriented network PSTN, during the transition between the networks the latter's line-oriented signaling ISUP is mapped onto the packet-oriented signaling SIP and SIP_T and its line-oriented bearer TDM is converted to the packet-oriented bearer RTP/RTCP_{A/IVR} (and vice versa). For example the SIP signaling SIP:Invite is mapped during interworking between the protocol ISUP and the protocol SIP onto the ISUP signaling O:IAM. The ISUP signalings O:ACM and O:ANM, ~~which indicate the ringing of the telephone T and acceptance of the call by the subscriber A,~~ are similarly mapped onto the SIP messages 180:Ringing and 200:OK. The first connection thus set up comprises at least one (in the case of a telephone call generally bi-directional) bearer TDM, RTP/RTCP_{IVR} for the transmission of information between the subscriber A and the system IVR.